

## US009761235B2

## (12) United States Patent

Wang et al.

# ENCODING METHOD, DECODING METHOD, ENCODING APPARATUS, AND

(71) Applicant: Huawei Technologies Co., Ltd.,

**DECODING APPARATUS** 

Shenzhen (CN)

(72) Inventors: Bin Wang, Beijing (CN); Zexin Liu,

Beijing (CN); Lei Miao, Beijing (CN)

(73) Assignee: HUAWEI TECHNOLOGIES CO.,

LTD., Shenzhen (CN)

(\*) Notice: Subject to any disclaimer, the term of this

patent is extended or adjusted under 35

U.S.C. 154(b) by 28 days.

(21) Appl. No.: 14/721,606

(22) Filed: May 26, 2015

(65) Prior Publication Data

US 2015/0255080 A1 Sep. 10, 2015

## Related U.S. Application Data

(63) Continuation of application No. PCT/CN2013/080061, filed on Jul. 25, 2013.

## (30) Foreign Application Priority Data

(51) **Int. Cl.** 

G10L 21/00 (2013.01) G10L 19/03 (2013.01)

(Continued)

(52) **U.S. Cl.** 

CPC ...... *G10L 19/03* (2013.01); *G10L 19/12* (2013.01); *G10L 19/26* (2013.01); *G10L 19/265* (2013.01); *G10L* 19/265 (2013.01);

## (Continued)

## (58) Field of Classification Search

USPC ....... 704/205, 211–213, 220, 258, 500–504 See application file for complete search history.

(10) Patent No.: US 9,761,235 B2

(45) **Date of Patent:** Sep. 12, 2017

## (56) References Cited

### U.S. PATENT DOCUMENTS

(Continued)

## FOREIGN PATENT DOCUMENTS

CN 1484756 A 3/2004 CN 101140759 A 3/2008 (Continued)

## OTHER PUBLICATIONS

Chen, J., et al., "Adaptive Postfiltering for Quality Enhancement of Coded Speech," IEEE Transactions on Speech and Audio Processing, vol. 3, No. 1, Jan. 1995, pp. 59-71.

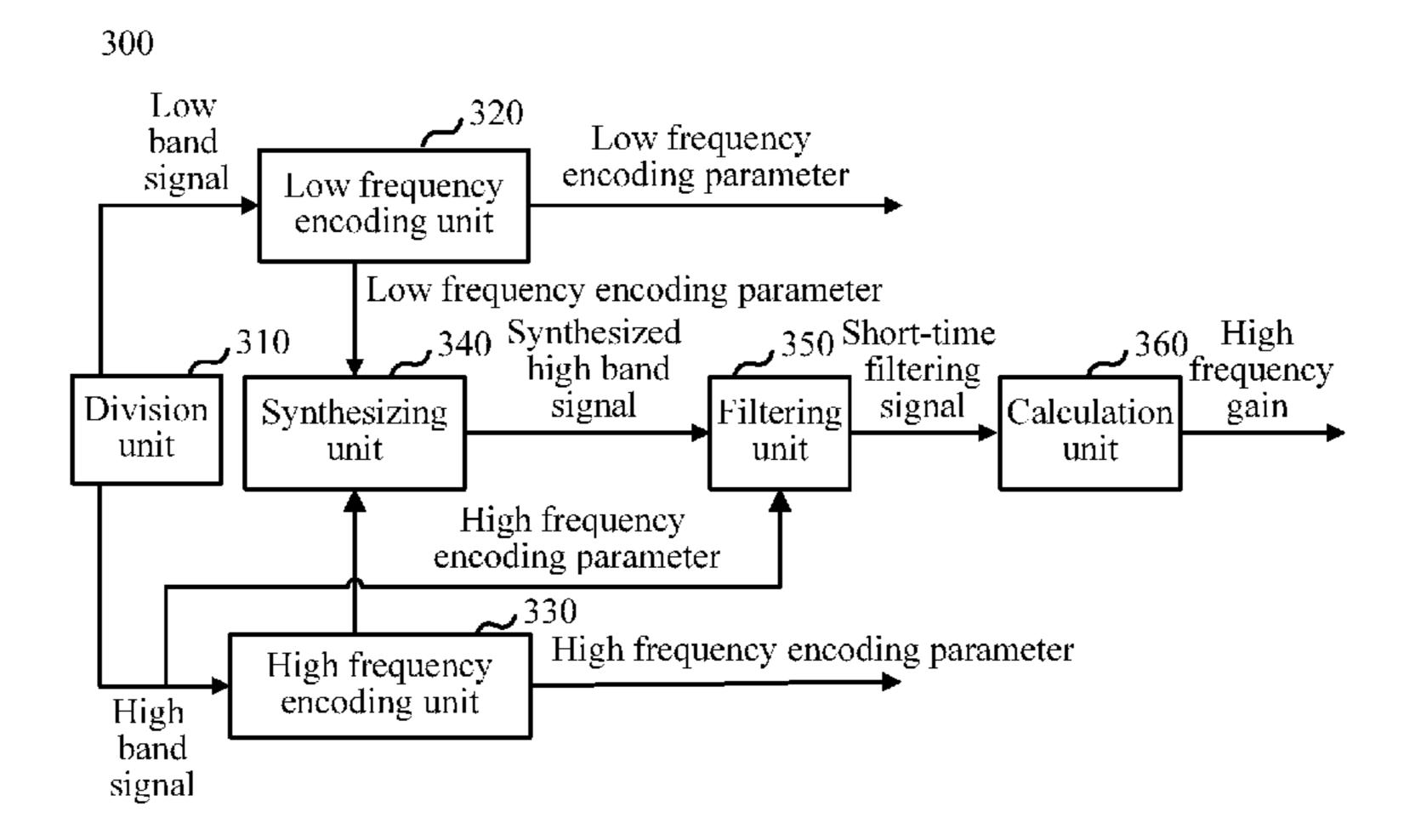
(Continued)

Primary Examiner — Leonard Saint Cyr (74) Attorney, Agent, or Firm — Conley Rose, P.C.

## (57) ABSTRACT

An encoding method, a decoding method, an encoding apparatus, a decoding apparatus, a transmitter, a receiver, and a communications system. The encoding method includes: dividing a to-be-encoded time-domain signal into a low band signal and a high band signal; performing encoding on the low band signal to obtain a low frequency encoding parameter; performing encoding on the high band signal to obtain a high frequency encoding parameter, and obtaining a synthesized high band signal; performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal; and calculating a high frequency gain based on the high band signal and the short-time filtering signal. A technical solution according to the embodiments of the present invention can improve an encoding and/or decoding effect.

## 18 Claims, 5 Drawing Sheets



(51)	Int. Cl.	
	G10L 21/038	(2013.01)
	G10L 19/26	(2013.01)
	G10L 19/12	(2013.01)
	G10L 19/02	(2013.01)
	G10L 19/00	(2013.01)
(52)	U.S. Cl.	
	CPC <i>G10L</i>	21/038 (2013.01); G10L 19/0204

(56)

## U.S. PATENT DOCUMENTS

**References Cited** 

(2013.01); G10L 2019/0016 (2013.01)

2003/0088423 A1	5/2003	Nishio et al.
2006/0271356 A1	11/2006	Vos
2008/0027718 A1	1/2008	Krishnan et al.
2008/0219344 A1	9/2008	Suzuki et al.
2009/0210234 A1	8/2009	Sung et al.
2009/0232228 A1*	9/2009	Thyssen G10L 19/005
		375/242
2009/0265167 A1*	10/2009	Ehara G10L 19/265
		704/219
2009/0319277 A1*	12/2009	Black G10L 21/038
		704/500
2011/0257984 A1*	10/2011	Virette G10L 19/26
		704/503
2011/0295598 A1*	12/2011	Yang G10L 21/038
		704/205
2012/0010882 A1*	1/2012	Thyssen G10L 19/005
		704/226

## FOREIGN PATENT DOCUMENTS

CN	101185124 A	5/2008
CN	101261834 A	9/2008
EP	2051245 A2	4/2009
JP	2008535024 A	8/2008
JP	2009545775 A	12/2009
KR	20070118173 A	12/2007
WO	9819298 A1	5/1998

WO	2006107840 A1	10/2006
WO	2006116025 A1	11/2006
WO	2013066238 A2	5/2013

## OTHER PUBLICATIONS

Foreign Communication From a Counterpart Application, European Application No. 13872123.8, Extended European Search Report dated Aug. 20, 2015, 7 pages.

Foreign Communication From A Counterpart Application, Chinese Application No. 201310014342.4, Chinese Office Action dated Jan. 7, 2016, 3 pages.

Foreign Communication From A Counterpart Application, Chinese Application No. 201310014342.4, Chinese Search Report dated Dec. 21, 2015, 2 pages.

Foreign Communication From A Counterpart Application, Korean Application No. 10-2015-7014971, Korean Office Action dated Dec. 16, 2015, 5 pages.

Foreign Communication From A Counterpart Application, Korean Application No. 10-2015-7014971, English Translation of Korean Office Action dated Dec. 16, 2015, 4 pages.

Foreign Communication From A Counterpart Application, PCT Application No. PCT/CN2013/080061, English Translation of International Search Report dated Nov. 7, 2013, 3 pages.

Foreign Communication From A Counterpart Application, PCT Application No. PCT/CN2013/080061, English Translation of Written Opinion dated Nov. 7, 2013, 9 pages.

Partial English Translation and Abstract of Japanese Patent Application No. JPA2008535024, Aug. 19, 2016, 42 pages.

Partial English Translation and Abstract of Japanese Patent Application No. JPA2009545775, Aug. 19, 2016, 79 pages.

Zhan, J., et al., "Bandwidth Extensions for China AVS-M Standard," IEEE International Conference on Acoustics, Spech and Signal Processing, Apr. 24, 2009, pp. 4149-4152.

Foreign Communication From A Counterpart Application, Japanese Application No. 2015-546810, Japanese Office Action dated Jul. 19, 2016, 3 pages.

Foreign Communication From A Counterpart Application, Japanese Application No. 2015-546810, English Translation of Japanese Office Action dated Jul. 19, 2016, 3 pages.

<sup>\*</sup> cited by examiner

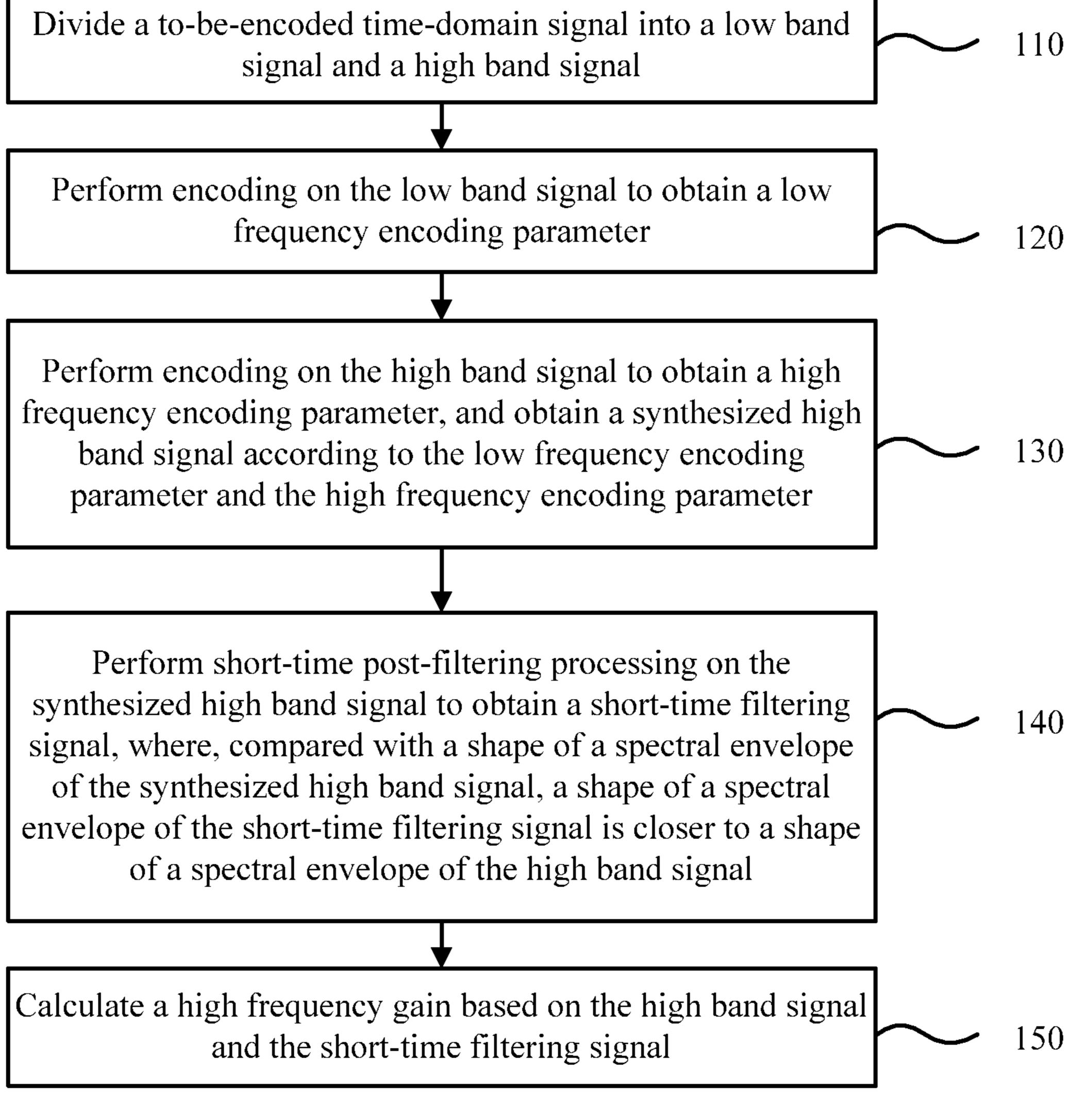


FIG. 1

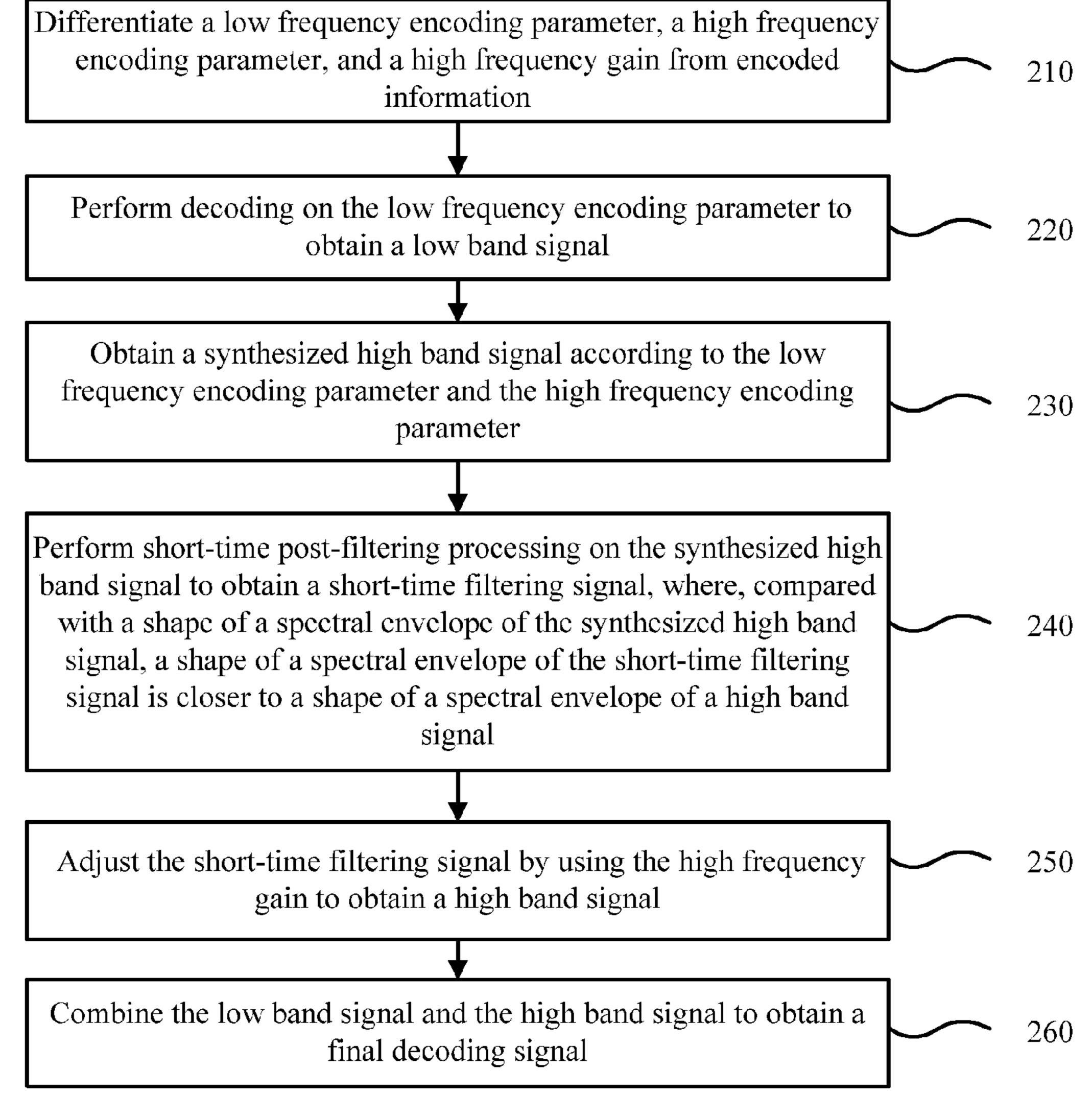


FIG. 2

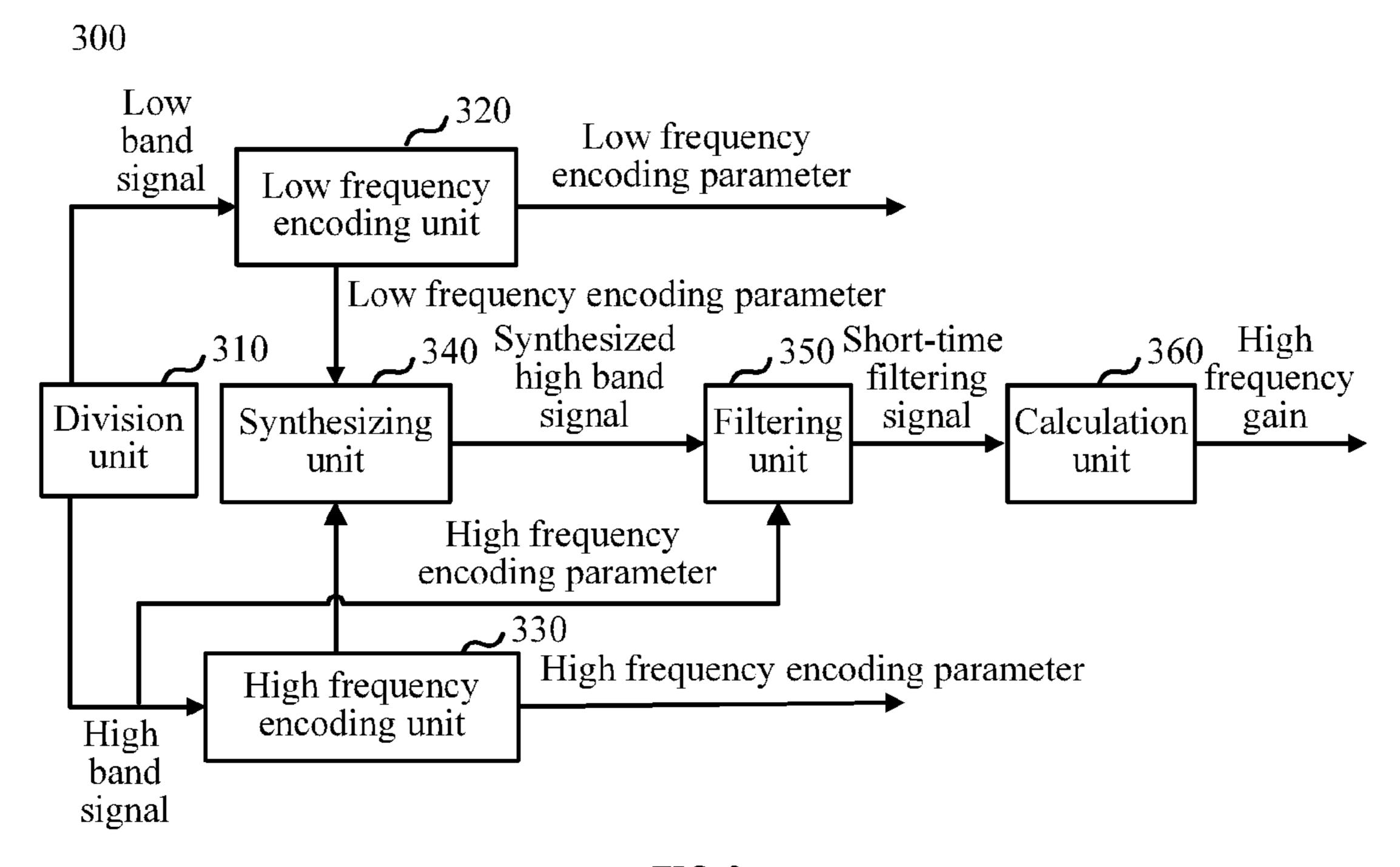


FIG. 3

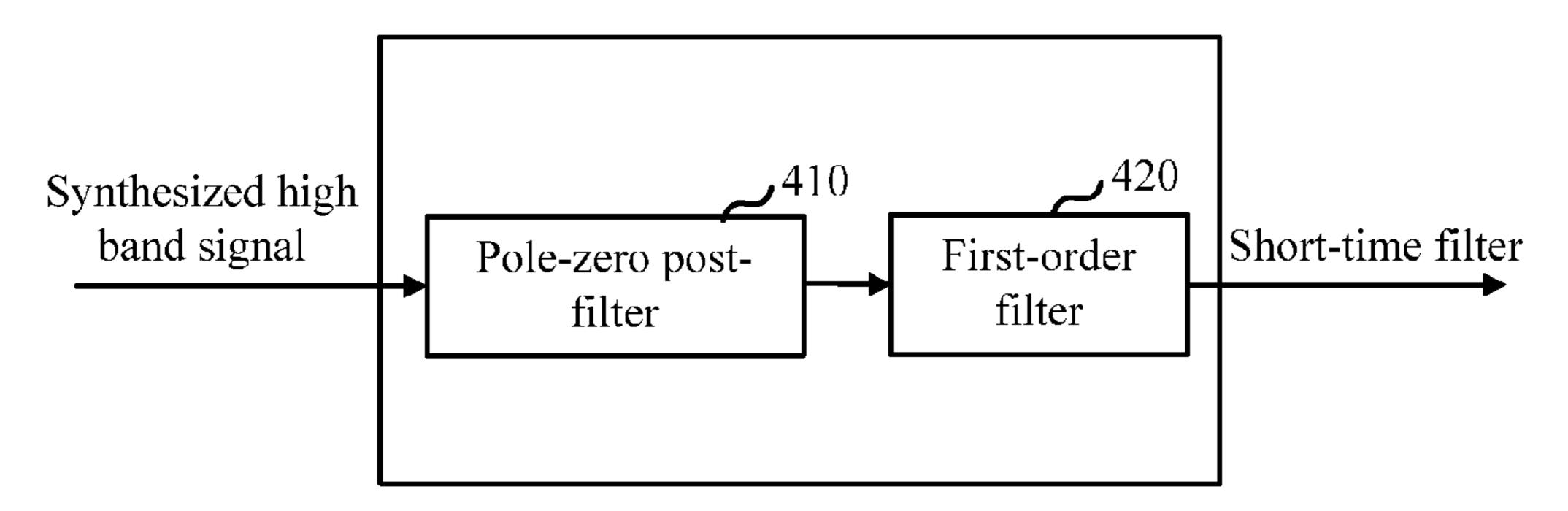


FIG. 4

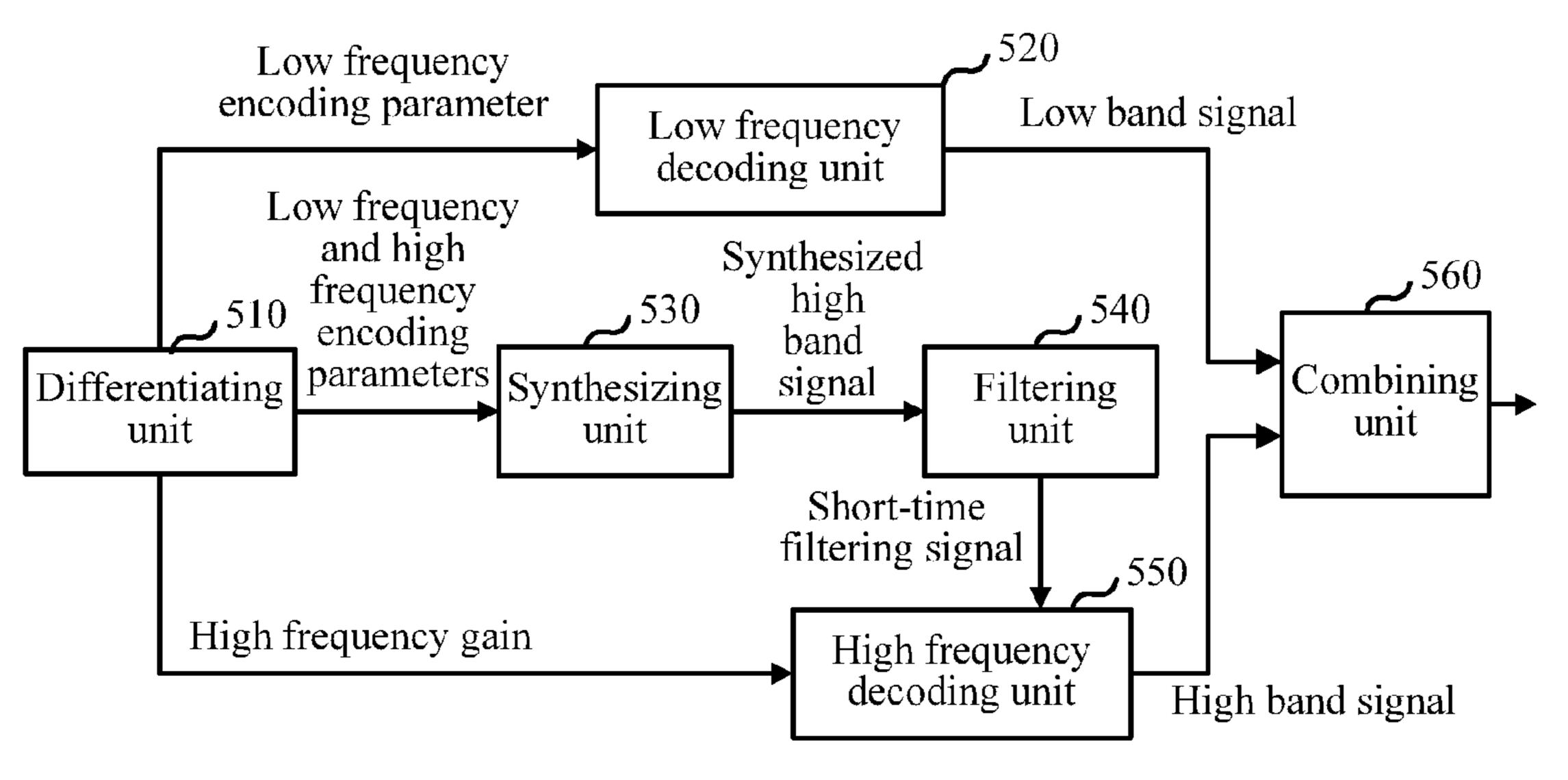
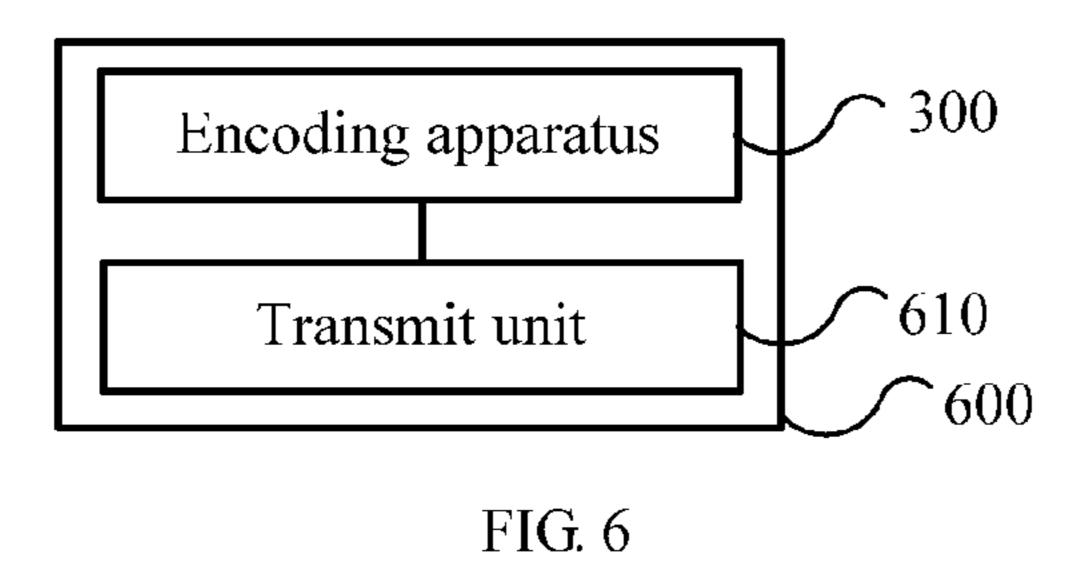


FIG. 5



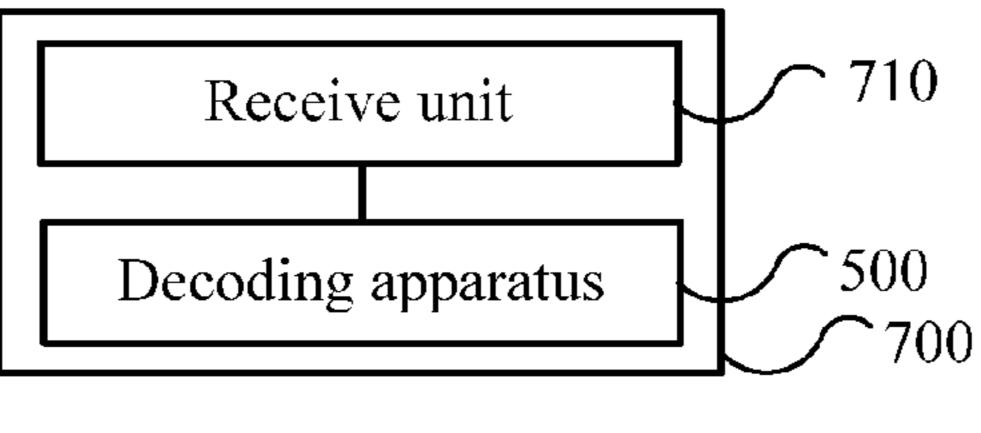


FIG. 7

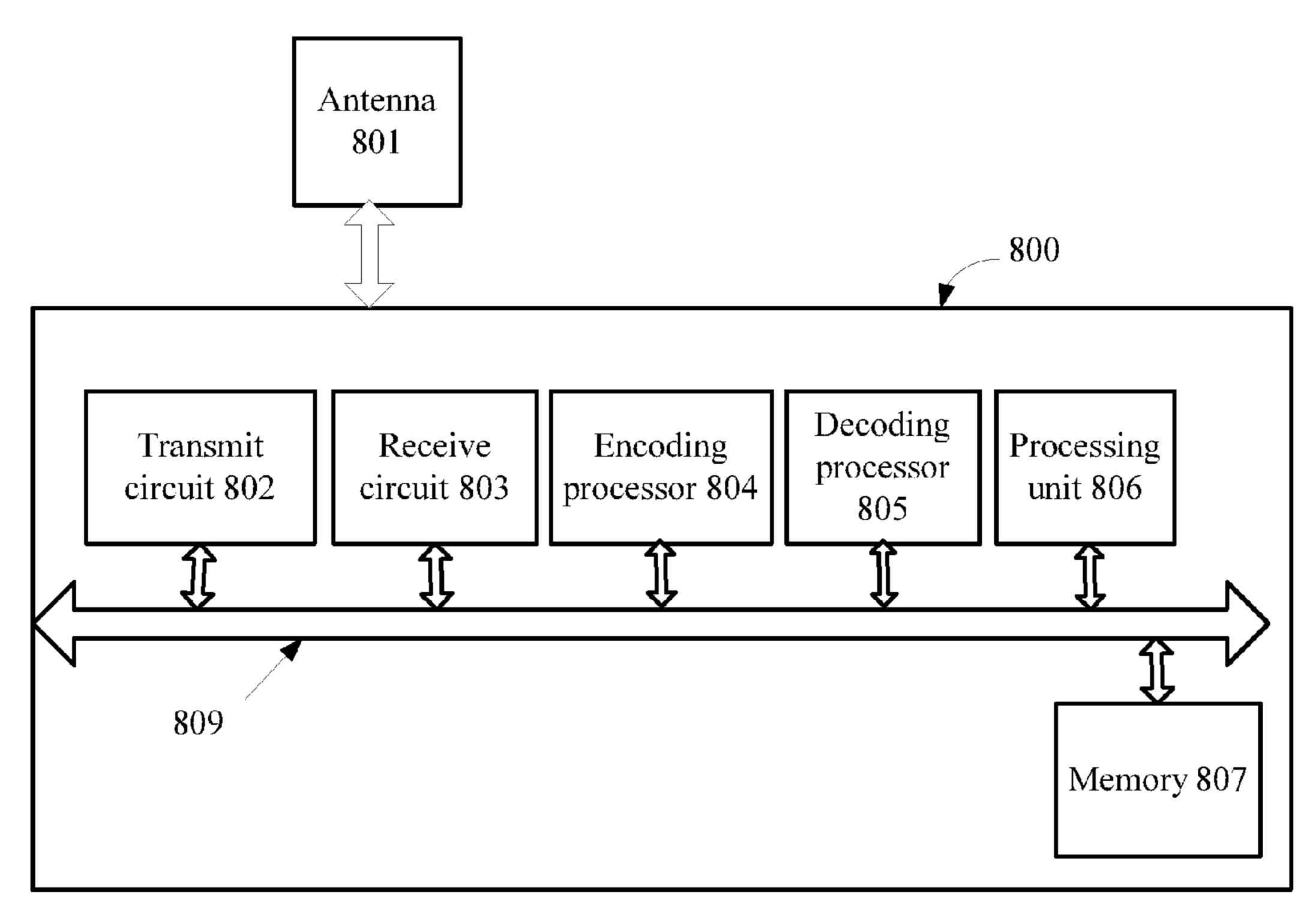


FIG. 8

# ENCODING METHOD, DECODING METHOD, ENCODING APPARATUS, AND DECODING APPARATUS

## CROSS-REFERENCE TO RELATED APPLICATION

This application is a continuation of International Application No. PCT/CN2013/080061, filed on Jul. 25, 2013, which claims priority to Chinese Patent Application No. 10 201310014342.4, filed on Jan. 15, 2013, both of which are hereby incorporated by reference in their entireties.

## TECHNICAL FIELD

Embodiments of the present invention relate to the field of communications technologies, and in particular, to an encoding method, a decoding method, an encoding apparatus, a decoding apparatus, a transmitter, a receiver, and a communications system.

## BACKGROUND

With continuous progress of communications technologies, users are imposing an increasingly high requirement on 25 voice quality. Generally, voice quality is improved by increasing bandwidth of the voice quality. If a signal whose bandwidth is wider is encoded in a traditional encoding manner, a bit rate is greatly improved and as a result, it is difficult to implement encoding because of a limitation 30 condition of current network bandwidth. Therefore, encoding needs to be performed on a signal whose bandwidth is wider in a case in which a bit rate is unchanged or slightly changed, and a solution proposed for this issue is to use a bandwidth extension technology. The bandwidth extension 35 technology may be completed in a time domain or a frequency domain. A basic principle of performing bandwidth extension in a time domain is that two different processing methods are used for a low band signal and a high band signal. For a low band signal in an original signal, encoding 40 is performed at an encoder side according to a requirement using various encoders; at a decoder side, a decoder corresponding to the encoder of the encoder side is used to decode and restore the low band signal. For a high band signal, at the encoder side, an encoder used for the low band signal is 45 used to obtain a low frequency encoding parameter so as to predict a high frequency excitation signal, processing is performed on a high band signal in an original signal to obtain a high frequency encoding parameter, and a synthesized high band signal is obtained based on the high fre- 50 quency encoding parameter and the high frequency excitation signal; then the synthesized high band signal and the high band signal in the original signal are compared to obtain a high frequency gain that is used to adjust a gain of the high band signal, and the high frequency gain and the 55 high frequency encoding parameter are transferred to the decoder side to restore the high band signal. At the decoder side, the low frequency encoding parameter that is extracted when the low band signal is decoded is used to restore the high frequency excitation signal, the synthesized high band 60 signal is obtained based on the high frequency excitation signal and the high frequency encoding parameter that is extracted when the high band signal is decoded, then a high frequency gain is adjusted for the synthesized high band signal to obtain a final high band signal, and the high band 65 signal and the low band signal are combined to obtain a final output signal.

2

In the foregoing technology of performing bandwidth extension in a time domain, the high band signal is restored in a condition of a specific rate, however, a performance indicator is deficient. It may be learned by comparing a frequency spectrum of a voice signal that is restored by decoding and a frequency spectrum of an original voice signal that, a restored voice signal sounds rustling and a sound is not clear enough.

## **SUMMARY**

Embodiments of the present invention provide an encoding method, a decoding method, an encoding apparatus, a decoding apparatus, a transmitter, a receiver, and a communications system, which can improve articulation of a restored signal, thereby enhancing encoding and decoding performance.

According to a first aspect, an encoding method is provided, including: dividing a to-be-encoded time-domain signal into a low band signal and a high band signal; 20 performing encoding on the low band signal to obtain a low frequency encoding parameter; performing encoding on the high band signal to obtain a high frequency encoding parameter, and obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal; and calculating a high frequency gain based on the high band signal and the short-time filtering signal.

With reference to the first aspect, in an implementation manner of the first aspect, the performing short-time postfiltering processing on the synthesized high band signal includes setting a coefficient of a pole-zero post-filter based on the high frequency encoding parameter, and performing filtering processing on the synthesized high band signal using the pole-zero post-filter.

With reference to the first aspect and the foregoing implementation manner, in another implementation manner of the first aspect, the performing short-time post-filtering processing on the synthesized high band signal may further include: after performing filtering processing on the synthesized high band signal using the pole-zero post-filter, performing, using a first-order filter whose z-domain transfer function is  $H_t(z)=1-\mu z^{-1}$ , filtering processing on the synthesized high band signal that has been processed by the pole-zero post-filter, where  $\mu$  is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

With reference to the first aspect and the foregoing implementation manners, in another implementation manner of the first aspect, the performing encoding on the high band signal to obtain a high frequency encoding parameter includes performing, using a linear predictive coding LPC technology, encoding on the high band signal to obtain an LPC coefficient and use the LPC coefficient as the high frequency encoding parameter, where a z-domain transfer function of the pole-zero post-filter is a formula as follows:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$

where  $\alpha_1, \alpha_2, \dots \alpha_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

With reference to the first aspect and the foregoing implementation manners, in another implementation manner <sup>5</sup> of the first aspect, the encoding method may further include generating an encoding bitstream according to the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain.

According to a second aspect, a decoding method is 10 provided, including: differentiating a low frequency encoding parameter, a high frequency encoding parameter, and a high frequency gain from encoded information; performing decoding on the low frequency encoding parameter to obtain a low band signal; obtaining a synthesized high band signal 15 according to the low frequency encoding parameter and the high frequency encoding parameter; performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band <sup>20</sup> signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of a high band signal; adjusting the short-time filtering signal using the high frequency gain to obtain a high band signal; and combining the low band signal and the high band signal <sup>25</sup> to obtain a final decoding signal.

With reference to the second aspect, in an implementation manner of the second aspect, the performing short-time post-filtering processing on the synthesized high band signal includes: setting a coefficient of a pole-zero post-filter based <sup>30</sup> on the high frequency encoding parameter, and performing filtering processing on the synthesized high band signal using the pole-zero post-filter.

With reference to the second aspect and the foregoing implementation manner, in another implementation manner <sup>35</sup> of the second aspect, the performing short-time post-filtering processing on the synthesized high band signal may further include: after performing filtering processing on the synthesized high band signal using the pole-zero post-filter, performing, using a first-order filter whose z-domain transfer 40 function is  $H_z(z)=1-\mu z^{-1}$ , filtering processing on the synthesized high band signal that has been processed by the pole-zero post-filter, where μ is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

With reference to the second aspect and the foregoing implementation manners, in another implementation manner of the second aspect, the high frequency encoding parameter may include an LPC coefficient that is obtained by performing encoding using a linear predictive coding LPC technology, and a z-domain transfer function of the pole-zero post-filter is a formula as follows:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$

of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

According to a third aspect, an encoding apparatus is provided, including: a division unit configured to divide a to-be-encoded time-domain signal into a low band signal 65 and a high band signal; a low frequency encoding unit configured to perform encoding on the low band signal to

obtain a low frequency encoding parameter; a high frequency encoding unit configured to perform encoding on the high band signal to obtain a high frequency encoding parameter; a synthesizing unit configured to obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; a filtering unit configured to perform short-time postfiltering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal; and a calculation unit configured to calculate a high frequency gain based on the high band signal and the short-time filtering signal.

With reference to the third aspect, in an implementation manner of the third aspect, the filtering unit may include a pole-zero post-filter configured to perform filtering processing on the synthesized high band signal, where a coefficient of the pole-zero post-filter may be set based on the high frequency encoding parameter.

With reference to the third aspect and the foregoing implementation manner, in another implementation manner of the third aspect, the filtering unit may further include a first-order filter, which is located behind the pole-zero post-filter and whose z-domain transfer function is  $H_r(z)=1$ μz<sup>-1</sup> configured to perform filtering processing on the synthesized high band signal that has been processed by the pole-zero post-filter, where μ is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

With reference to the third aspect and the foregoing implementation manners, in another implementation manner of the third aspect, the high frequency encoding unit may perform encoding on the high band signal using a linear predictive coding (LPC) technology to obtain an LPC coefficient and use the LPC coefficient as the high frequency encoding parameter, and a z-domain transfer function of the pole-zero post-filter is a formula as follows:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$

where  $\alpha_1, \alpha_2, \dots \alpha_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

With reference to the third aspect and the foregoing implementation manners, in another implementation manner of the third aspect, the encoding apparatus may further include a bitstream generating unit configured to generate an encoding bitstream according to the low frequency encoding 55 parameter, the high frequency encoding parameter, and the high frequency gain.

According to a fourth aspect, a decoding apparatus is provided, including: a differentiating unit configured to differentiate a low frequency encoding parameter, a high where  $\alpha_1, \alpha_2, \dots \alpha_M$  is the LPC coefficient, M is an order 60 frequency encoding parameter, and a high frequency gain from encoded information; a low frequency decoding unit configured to perform decoding on the low frequency encoding parameter to obtain a low band signal; a synthesizing unit configured to obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; a filtering unit configured to perform short-time post-filtering processing on the

synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of a high band signal; a high frequency decoding unit configured to adjust the short-time filtering signal using the high frequency gain to obtain a high band signal; and a combining unit configured to combine the low band signal and the high band signal to obtain a final decoding signal.

With reference to the fourth aspect, in an implementation manner of the fourth aspect, the filtering unit may include a pole-zero post-filter configured to perform filtering processing on the synthesized high band signal, where a coefficient of the pole-zero post-filter may be set based on the high frequency encoding parameter.

With reference to the fourth aspect and the foregoing implementation manner, in another implementation manner of the fourth aspect, the filtering unit may further include a 20 first-order filter, which is located behind the pole-zero post-filter and whose z-domain transfer function is  $H_t(z)=1-\mu z^{-1}$  configured to perform filtering processing on the synthesized high band signal that has been processed by the pole-zero post-filter, where  $\mu$  is a preset constant or a value 25 obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

With reference to the fourth aspect and the foregoing implementation manners, in another implementation manner of the fourth aspect, the high frequency encoding parameter may include an LPC coefficient that is obtained using an LPC technology, and a z-domain transfer function of the pole-zero post-filter is a formula as follows:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$

where  $\alpha_1, \alpha_2, \dots \alpha_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

According to a fifth aspect, a transmitter is provided, including an encoding apparatus according to the third aspect, and a transmit unit configured to allocate bits to a high frequency encoding parameter and a low frequency encoding parameter that are generated by the encoding apparatus so as to generate a bit stream, and transmit the bit stream.

creative efforts shared present invention.

The technical so applied to various System for Mobile Multiple Access (Westermann) tiple Access (Westermann).

According to a sixth aspect, a receiver is provided, including a receive unit configured to receive a bit stream and extract encoded information from the bit stream; and a decoding apparatus according to the fourth aspect.

According to a seventh aspect, a communications system 55 domain. is provided, including a transmitter according the fifth aspect FIG. 3 or a receiver according to the sixth aspect.

In the foregoing technical solution according to the embodiments of the present invention, when a high frequency gain is calculated based on a synthesized high band 60 signal in an encoding and decoding process, short-time post-filtering processing is performed on the synthesized high band signal to obtain a short-time filtering signal, and the high frequency gain is calculated based on the short-time filtering signal, which can reduce or even remove a rustle 65 from a restored signal, and improve an encoding and decoding effect.

6

## BRIEF DESCRIPTION OF DRAWINGS

To describe the technical solutions in the embodiments of the present invention more clearly, the following briefly introduces the accompanying drawings required for describing the embodiments or the prior art. Apparently, the accompanying drawings in the following description show merely some embodiments of the present invention, and a person of ordinary skill in the art may still derive other drawings from these accompanying drawings without creative efforts.

FIG. 1 is a flowchart that schematically shows an encoding method according to an embodiment of the present invention;

ing on the synthesized high band signal, where a coefficient of the pole-zero post-filter may be set based on the high

FIG. 3 is a block diagram that schematically shows an encoding apparatus according to an embodiment of the present invention;

FIG. 4 is a block diagram that schematically shows a filtering unit in an encoding apparatus according to an embodiment of the present invention;

FIG. 5 is a block diagram that schematically shows a decoding apparatus according to an embodiment of the present invention;

FIG. 6 is a block diagram that schematically shows a transmitter according to an embodiment of the present invention;

FIG. 7 is a block diagram that schematically shows a receiver according to an embodiment of the present invention; and

FIG. 8 is a schematic block diagram of an apparatus according to another embodiment of the present invention.

## DESCRIPTION OF EMBODIMENTS

The following clearly describes the technical solutions in the embodiments of the present invention with reference to the accompanying drawings in the embodiments of the present invention. The described embodiments are some but not all of the embodiments of the present invention. All other embodiments obtained by a person of ordinary skill in the art based on the embodiments of the present invention without creative efforts shall fall within the protection scope of the present invention.

The technical solutions of the present invention may be applied to various communications systems, such as Global System for Mobile Communication (GSM), Code Division Multiple Access (CDMA), Wideband Code Division Multiple Access (WCDMA), general packet radio service (GPRS), and Long Term Evolution (LTE).

A bandwidth extension technology may be completed in a time domain or a frequency domain, and in the present invention, bandwidth extension is completed in a time domain.

FIG. 1 is a flowchart that schematically shows an encoding method 100 according to an embodiment of the present invention. The encoding method 100 includes: dividing a to-be-encoded time-domain signal into a low band signal and a high band signal (110); performing encoding on the low band signal to obtain a low frequency encoding parameter (120); performing encoding on the high band signal to obtain a high frequency encoding parameter, and obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter (130); performing short-time post-filtering processing on the synthesized high band signal to obtain a

short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal (140); and calculating a high frequency gain based on 5 the high band signal and the short-time filtering signal (150).

In 110, the to-be-encoded time-domain signal is divided into the low band signal and the high band signal. This division is to divide the time-domain signal into two signals for processing, so that the low band signal and the high band 10 signal can be separately processed. The division may be implemented using any conventional or future division technology. The meaning of the low frequency herein is relative to the meaning of the high frequency. For example, a frequency threshold may be set, where a frequency lower 15 than the frequency threshold is a low frequency, and a frequency higher than the frequency threshold is a high frequency. In practice, the frequency threshold may be set according to a requirement, and a low band signal component and a high frequency component in a signal may also 20 be differentiated using another manner, so as to implement the division.

In 120, the low band signal is encoded to obtain the low frequency encoding parameter. By the encoding, the low band signal is processed so as to obtain the low frequency 25 encoding parameter, so that a decoder side restores the low band signal according to the low frequency encoding parameter. The low frequency encoding parameter is a parameter required by the decoder side to restore the low band signal. As an example, encoding may be performed using an 30 encoder (Algebraic Code Excited Linear Prediction (ACELP) encoder) that uses an ACELP algorithm, and a low frequency encoding parameter obtained in this case may include, for example, an algebraic codebook, an algebraic codebook gain, an adaptive codebook, an adaptive codebook 35 gain, and a pitch period, and may also include another parameter. The low frequency encoding parameter may be transferred to the decoder side to restore the low band signal. In addition, when the algebraic codebook and the adaptive codebook are transferred from an encoder side to the 40 decoder side, only an algebraic codebook index and an adaptive codebook index may be transferred, and the decoder side obtains a corresponding algebraic codebook and adaptive codebook according to the algebraic codebook index and the adaptive codebook index, so as to implement 45 the restoration. In practice, the low band signal may be encoded using a proper encoding technology according to a requirement. When an encoding technology changes, composition of the low frequency encoding parameter may also change.

In this embodiment of the present invention, an encoding technology that uses the ACELP algorithm is used as an example for description.

In 130, the high band signal is encoded to obtain the high frequency encoding parameter, and the synthesized high 55 band signal is obtained according to the low frequency encoding parameter and the high frequency encoding parameter. For example, linear predictive coding (LPC) analysis may be performed on a high band signal in an original signal to obtain a high frequency encoding parameter such as an 60 LPC coefficient, the low frequency encoding parameter is used to predict a high frequency excitation signal, and the high frequency excitation signal is used to obtain the synthesized high band signal using a synthesis filter that is determined according to the LPC coefficient. In practice, 65 another technology may be adopted according to a requirement so as to obtain the synthesized high band signal

8

according to the low frequency encoding parameter and the high frequency encoding parameter.

In a process of obtaining the synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter, a frequency spectrum of the high frequency excitation signal that is obtained using the low frequency encoding parameter to perform a prediction is flat; however, a frequency spectrum of an actual high frequency excitation signal is not flat. This difference causes that the spectral envelope of the synthesized high band signal does not change with the spectral envelope of the high band signal in the original signal, and further causes a rustle in a restored voice signal.

In 140, the short-time post-filtering processing is performed on the synthesized high band signal to obtain the short-time filtering signal, where, compared with the shape of the spectral envelope of the synthesized high band signal, the shape of the spectral envelope of the short-time filtering signal is closer to the shape of the spectral envelope of the high band signal.

For example, a filter that is used to perform post-filtering processing on the synthesized high band signal may be formed based on the high frequency encoding parameter, and the filter is used to perform filtering on the synthesized high band signal to obtain the short-time filtering signal, where, compared with the shape of the spectral envelope of the synthesized high band signal, the shape of the spectral envelope of the short-time filtering signal is closer to the shape of the spectral envelope of the high band signal. For example, a coefficient of a pole-zero post-filter may be set based on the high frequency encoding parameter, and the pole-zero post-filter may be used to perform filtering processing on the synthesized high band signal. Alternatively, a coefficient of an all-pole post-filter may be set based on the high frequency encoding parameter, and the all-pole postfilter may be used to perform filtering processing on the synthesized high band signal. That encoding is performed on the high band signal using an LPC technology is used as an example for description below.

In a case in which encoding is performed on the high band signal using the LPC technology, the high frequency encoding parameter includes an LPC coefficient  $\alpha_1, \alpha_2, \ldots, \alpha_M$ , is an order of the LPC coefficient, and a pole-zero post-filter whose coefficient transfer function is calculated in the following formula (1) may be set based on the LPC coefficient:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$
formula (1)

where  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ . In practice, it may be made that  $\beta = 0.5$ ,  $\gamma = 0.8$ . A shape of a spectral envelope of a synthesized high band signal that has been processed by the pole-zero post-filter whose transfer function is shown in formula (1) is closer to the shape of the spectral envelope of the high band signal, so as to avoid a rustle in the restored signal and improve an encoding effect. The transfer function shown in formula (1) is a z-domain transfer function, but this transfer function may further be a transfer function in another domain such as a time domain or a frequency domain.

In addition, the synthesized high band signal after the pole-zero post-filtering processing has a low-pass effect, therefore, after the filtering processing is performed on the synthesized high band signal using the pole-zero post-filter,

processing may further be performed using a first-order filter whose z-domain transfer function is calculated in the following formula (2):

$$H_t(z)=1-\mu z^{-1}$$
 formula (2) 5

where  $\mu$  is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal. For example, in a case in which encoding is performed on the high band signal using the LPC technology, μ may be obtained by calculation using the LPC coefficient,  $\beta$  and  $\gamma$ , and the synthesized high band signal as a function, and a person skilled in the art may use various existing methods to perform the calculation, and details are not described herein again. Compared with a short-time filtering signal that is obtained from filtering processing only by the pole-zero post-filter, a change of a spectral envelope of a short-time filtering signal that is obtained from filtering processing by both the pole-zero post-filter and the firstorder filter is closer to a change of the spectral envelope of the original high band signal, and an encoding effect can be further improved.

In a case in which encoding is performed on the high band signal using the LPC technology, if the short-time postfiltering processing is implemented using the all-pole postfilter, a z-domain transfer function of the all-pole post-filter whose coefficient is set based on the high frequency encoding parameter may be shown in the following formula (3):

$$H_s(z) = \frac{1}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}}$$
 formula (3)

 $\alpha_1, \alpha_2, \ldots \alpha_M$  is used as an LPC coefficient of the high frequency encoding parameter, and M is an order of the LPC coefficient.

In 150, the high frequency gain is calculated based on the high band signal and the short-time filtering signal. The high 40 frequency gain is used to indicate an energy difference between the original high band signal and the short-time filtering signal (that is, a synthesized high band signal after short-time post-filtering processing). When signal decoding is performed, after the synthesized high band signal is 45 obtained, the high frequency gain can be used to restore a high band signal.

After the high frequency gain, the high frequency encoding parameter, and the low frequency encoding parameter are obtained, an encoding bitstream is generated according 50 to the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain, thereby implementing encoding. In the foregoing encoding method according to this embodiment of the present invention, short-time post-filtering processing is performed on a syn- 55 thesized high band signal to obtain a short-time filtering signal, and a high frequency gain is calculated based on the short-time filtering signal, which can reduce or even remove a rustle from a restored signal, and improve an encoding effect.

FIG. 2 is a flowchart that schematically shows a decoding method 200 according to an embodiment of the present invention. The decoding method 200 includes: differentiating a low frequency encoding parameter, a high frequency encoded information (210); performing decoding on the low frequency encoding parameter to obtain a low band signal

**10** 

(220); obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter (230); performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of a high band signal (240); adjusting the short-time filtering signal using the high frequency gain to obtain a high band signal (250); and combining the low band signal and the high band signal to obtain a final decoding signal (260).

In 210, the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain are differentiated from the encoded information. The low frequency encoding parameter may include, for example, an algebraic codebook, an algebraic codebook gain, an adaptive codebook, an adaptive codebook gain, a pitch period, and another parameter, and the high frequency encoding parameter may include, for example, an LPC coefficient and another parameter. In addition, the low frequency encoding parameter and the high frequency encoding parameter may alternatively include another parameter according to a different encoding technology.

In 220, decoding is performed on the low frequency encoding parameter to obtain the low band signal. A specific decoding manner corresponds to an encoding manner of an encoder side. For example, when an ACELP encoder that uses an ACELP algorithm is used at the encoder side to perform encoding, in 220, an ACELP decoder is used to obtain the low band signal.

In 230, the synthesized high band signal is obtained according to the low frequency encoding parameter and the high frequency encoding parameter. For example, the low where  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ , 35 frequency encoding parameter is used to restore a high frequency excitation signal, the LPC coefficient in the high frequency encoding parameter is used to generate a synthesized filter, and the synthesized filter is used to perform filtering on the high frequency excitation signal to obtain the synthesized high band signal. In practice, another technology may further be adopted according to a requirement so as to obtain the synthesized high band signal based on the low frequency encoding parameter and the high frequency encoding parameter.

> As described above, in a process of obtaining the synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter, a frequency spectrum of the high frequency excitation signal that is obtained using the low frequency encoding parameter to perform a prediction is flat, however, a frequency spectrum of an actual high frequency excitation signal is not flat. This difference causes that the spectral envelope of the synthesized high band signal does not change with a spectral envelope of the high band signal in an original signal, and further causes a rustle in a restored voice signal.

In 240, the short-time post-filtering processing is performed on the synthesized high band signal to obtain the short-time filtering signal, where, compared with the shape of the spectral envelope of the synthesized high band signal, the shape of the spectral envelope of the short-time filtering signal is closer to the shape of the spectral envelope of the high band signal.

For example, a filter that is used to perform post-filtering encoding parameter, and a high frequency gain from 65 processing on the synthesized high band signal may be formed based on the high frequency encoding parameter, and the filter is used to perform filtering on the synthesized

high band signal to obtain a short-time filtering signal, where, compared with the synthesized high band signal, the shape of the spectral envelope of the short-time filtering signal is closer to the shape of the spectral envelope of the high band signal. For example, a coefficient of a pole-zero post-filter may be set based on the high frequency encoding parameter, and the pole-zero post-filter may be used to perform filtering processing on the synthesized high band signal. Alternatively, a coefficient of an all-pole post-filter may be set based on the high frequency encoding parameter, and the all-pole post-filter may be used to perform filtering processing on the synthesized high band signal.

In a case in which encoding is performed on the high band signal using an LPC technology, the high frequency encoding parameter includes an LPC coefficient  $\alpha_1, \alpha_2, \ldots \alpha_M$ , 15 M is an order of the LPC coefficient, a z-domain transfer function of a pole-zero post-filter that is set based on the LPC coefficient may be the foregoing formula (1), and a z-domain transfer function of an all-pole post-filter that is set based on the LPC coefficient may be the foregoing formula 20 (3). Compared with a shape of a spectral envelope of a synthesized high band signal that has not been processed by the pole-zero post-filter (or the all-pole post-filter), a shape of a spectral envelope of a synthesized high band signal that has been processed by the pole-zero post-filter (or the 25 all-pole post-filter) is closer to a shape of a spectral envelope of an original high band signal, which avoids a rustle in a restored signal, thereby improving an encoding effect.

In addition, as described above, the synthesized high band signal after the pole-zero post-filtering processing shown in 30 formula (1) has a low-pass effect, therefore, after the filtering processing is performed on the synthesized high band signal using the pole-zero post-filter, processing may further be performed using a first-order filter whose z-domain transfer function is the foregoing formula (2), so as to further 35 improve the encoding effect.

For description of 240, reference may be made to the foregoing description that is of 140 and is performed with reference to FIG. 1.

In **250**, the high frequency gain is used to adjust the 40 short-time filtering signal to obtain the high band signal. Corresponding to that, at the decoder side, the high frequency gain is obtained using the high band signal and the short-time filtering signal (**150** in FIG. **1**), in **250**, the high frequency gain is used to adjust the short-time filtering 45 signal to restore the high band signal.

In 260, the low band signal and the high band signal are combined to obtain the final decoding signal (260). This combination manner corresponds to a dividing manner in 110 of FIG. 1, thereby implementing decoding to obtain a 50 final output signal.

In the foregoing decoding method according to this embodiment of the present invention, short-time post-filtering processing is performed on a synthesized high band signal to obtain a short-time filtering signal, and a high 55 frequency gain is calculated based on the short-time filtering signal, which can reduce or even remove a rustle from a restored signal, and improve a decoding effect.

FIG. 3 is block diagram that schematically shows an encoding apparatus 300 according to an embodiment of the 60 present invention. The encoding apparatus 300 includes: a division unit 310 configured to divide a to-be-encoded time-domain signal into a low band signal and a high band signal; a low frequency encoding unit 320 configured to perform encoding on the low band signal to obtain a low 65 frequency encoding parameter; a high frequency encoding unit 330 configured to perform encoding on the high band

12

signal to obtain a high frequency encoding parameter; a synthesizing unit 340 configured to obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; a filtering unit 350 configured to perform short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal; and a calculation unit 360 configured to calculate a high frequency gain based on the high band signal and the short-time filtering signal.

After receiving an input time-domain signal, the division unit 310 divides the to-be-encoded time-domain signal into two signals (a low band signal and a high band signal) to perform processing. The division may be implemented using any conventional or future division technology. The meaning of the low frequency herein is relative to the meaning of the high frequency. For example, a frequency threshold may be set; where a frequency lower than the frequency threshold is a low frequency, and a frequency higher than the frequency threshold may be set according to a requirement, and a low band signal component and a high frequency component in a signal may also be differentiated using another manner, so as to implement the division.

The low frequency encoding unit 320 may use a proper encoding technology according to a requirement so as to perform encoding on the low band signal. For example, the low frequency encoding unit 320 may use an ACELP encoder to perform encoding so as to obtain the low frequency encoding parameter (which may include, for example, an algebraic codebook, an algebraic codebook gain, an adaptive codebook, an adaptive codebook gain, and a pitch period). When a used encoding technology changes, composition of the low frequency encoding parameter may also change. The obtained low frequency encoding parameter is a parameter required for restoring the low band signal, and the obtained low frequency encoding parameter is transferred to a decoder to restore the low band signal.

The high frequency encoding unit 330 performs encoding on the high band signal to obtain a high frequency encoding parameter. For example, the high frequency encoding unit 330 may perform LPC analysis on a high band signal in an original signal to obtain a high frequency encoding parameter such as an LPC coefficient. An encoding technology that is used to perform encoding on the high band signal constitutes no limitation on the embodiments of the present invention.

The synthesizing unit 340 uses the low frequency encoding parameter to predict a high frequency excitation signal, and enables the high frequency excitation signal to pass to a synthesized filter that is determined according to the LPC coefficient so as to obtain the synthesized high band signal. In practice, another technology may further be adopted according to a requirement so as to obtain the synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter. A frequency spectrum of the high frequency excitation signal that is obtained by the synthesizing unit 340 by performing a prediction using the low frequency encoding parameter is flat; however, a frequency spectrum of an actual high frequency excitation signal is not flat. This difference causes that the spectral envelope of the synthesized high band

signal does not change with the spectral envelope of the high band signal in the original signal, and further causes a rustle in a restored voice signal.

The filtering unit 350 is configured to perform short-time post-filtering processing on the synthesized high band signal 5 to obtain the short-time filtering signal, where, compared with the shape of the spectral envelope of the synthesized high band signal, the shape of the spectral envelope of the short-time filtering signal is closer to the shape of the spectral envelope of the high band signal. The following 10 describes the filtering unit 350 with reference to FIG. 4.

FIG. 4 is a block diagram that schematically shows the filtering unit 350 in the encoding apparatus 300 according to an embodiment of the present invention.

The filtering unit **350** may include a pole-zero post-filter 15 410, which is configured to perform filtering processing on the synthesized high band signal, where a coefficient of the pole-zero post-filter may be set based on the high frequency encoding parameter. In a case in which the high frequency encoding unit 330 performs encoding on the high band 20 signal using an LPC technology, a z-domain transfer function of the pole-zero post-filter 410 may be shown in the foregoing formula (1). A shape of a spectral envelope of the synthesized high band signal that is processed by the polezero post-filter 410 is closer to the shape of the spectral 25 envelope of the original high band signal, which avoids a rustle in a restored signal, thereby improving an encoding effect. Optionally, the filtering unit 350 may further include a first-order filter 420, which is located behind the pole-zero post-filter. A z-domain transfer function of the first-order 30 filter **420** may be shown in the foregoing formula (2). Compared with a short-time filtering signal that is obtained from filtering processing by the pole-zero post-filter 410 only, a change of a spectral envelope of a short-time filtering signal that is obtained from filtering processing by both the 35 pole-zero post-filter 410 and the first-order filter 420 is closer to a change of the spectral envelope of the original high band signal, and an encoding effect can be further improved.

As a replacement of the filtering unit **350** shown in FIG. **40 4**, an all-pole post-filter may further be used to perform short-time post-filtering processing to obtain the short-time filtering signal, where, compared with the shape of the spectral envelope of the synthesized high band signal, the shape of the spectral envelope of the short-time filtering 45 signal is closer to the shape of the spectral envelope of the high band signal. In a case in which encoding is performed on the high band signal using the LPC technology, a z-domain transfer function of the all-pole post-filter may be shown in the foregoing formula (3).

For description of the filtering unit 350, reference may be made to the foregoing description that is of 140 and is performed with reference to FIG. 1.

The calculation unit 360 calculates the high frequency gain based on the high band signal that is provided by the division unit and the short-time filtering signal that is output by the filtering unit 350. The high frequency gain and the low frequency encoding parameter and the high frequency encoding parameter together constitute encoding information, which is used for signal restoration at a decoder side.

In addition, the encoding apparatus 300 may further include a bitstream generating unit, where the bitstream generating unit is configured to generate an encoding bitstream according to the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain. The decoder side that receives the encoding bitstream may perform decoding based on the low frequency

14

encoding parameter, the high frequency encoding parameter, and the high frequency gain. For operations that are performed by units of the encoding apparatus shown in FIG. 3, reference may be made to the description that is of the encoding method and is performed with reference to FIG. 1.

In the foregoing encoding apparatus 300 according to this embodiment of the present invention, short-time post-filtering processing is performed on a synthesized high band signal to obtain a short-time filtering signal, and a high frequency gain is calculated based on the short-time filtering signal, which can reduce or even remove a rustle from a restored signal, and improve an encoding effect.

FIG. 5 is a block diagram that schematically shows a decoding apparatus 500 according to an embodiment of the present invention. The decoding apparatus 500 includes: a differentiating unit 510 configured to differentiate a low frequency encoding parameter, a high frequency encoding parameter, and a high frequency gain from encoded information; a low frequency decoding unit 520 configured to perform decoding on the low frequency encoding parameter to obtain a low band signal; a synthesizing unit 530 configured to obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; a filtering unit **540** configured to perform short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal; a high frequency decoding unit 550 configured to adjust the shorttime filtering signal using the high frequency gain to obtain a high band signal; and a combining unit 560 configured to combine the low band signal and the high band signal to obtain a final decoding signal.

The differentiating unit **510** differentiates the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain from encoded information. The low frequency encoding parameter may include, for example, an algebraic codebook, an algebraic codebook gain, an adaptive codebook, an adaptive codebook gain, a pitch period, and another parameter, and the high frequency encoding parameter may include, for example, an LPC coefficient and another parameter. In addition, the low frequency encoding parameter and the high frequency encoding parameter and the high frequency encoding parameter and alternatively include another parameter according to a different encoding technology.

The low frequency decoding unit **520** uses a decoding manner corresponding to an encoding manner of an encoder side, and performs decoding on the low frequency encoding parameter to obtain the low band signal. For example, when an ACELP encoder is used at the encoder side to perform encoding, the low frequency decoding unit **520** uses an ACELP decoder to obtain the low band signal.

That an LPC coefficient (that is, the high frequency encoding parameter) is obtained using LPC analysis is used as an example. The synthesizing unit **530** uses the low frequency encoding parameter to restore a high frequency excitation signal, uses the LPC coefficient to generate a synthesized filter, and uses the synthesized filter to perform filtering on the high frequency excitation signal to obtain the synthesized high band signal. In practice, another technology may further be adopted according to a requirement so as to obtain the synthesized high band signal based on the low frequency encoding parameter and the high frequency encoding parameter.

A frequency spectrum of the high frequency excitation signal that is obtained by the synthesizing unit **530** by performing a prediction using the low frequency encoding parameter is flat; however, a frequency spectrum of an actual high frequency excitation signal is not flat. This difference causes that the spectral envelope of the synthesized high band signal does not change with the spectral envelope of the high band signal in an original signal, and further causes a rustle in a restored voice signal.

For example, a structure of the filtering unit **540** may be shown in FIG. **4**. Alternatively, the filtering unit **540** may further use an all-pole post-filter to perform short-time post-filtering processing. In a case in which encoding is performed on the high band signal using an LPC technology, a z-domain transfer function of the all-pole post-filter may 15 be shown in the foregoing formula (3). The filtering unit **540** is the same as the filtering unit **350** in FIG. **3**; therefore, reference may be made to the foregoing description that is performed with reference to the filtering unit **350**.

Corresponding to an operation, in an encoding apparatus 20 300, of calculating a high frequency gain based on a high band signal and a short-time filtering signal, the high frequency decoding unit 550 uses the high frequency gain to adjust the short-time filtering signal so as to obtain the high band signal.

In a combining manner corresponding to a dividing manner used by the division unit in the encoding apparatus 300, the combining unit 560 combines the low band signal and the high band signal, thereby implementing decoding and obtaining a final output signal.

In the foregoing decoding apparatus **500** according to this embodiment of the present invention, short-time post-filtering processing is performed on a synthesized high band signal to obtain a short-time filtering signal, and a high frequency gain is calculated based on the short-time filtering signal, which can reduce or even remove a rustle from a restored signal, and improve a decoding effect.

FIG. 6 is a diagram block that schematically shows a transmitter 600 according to an embodiment of the present invention. The transmitter 600 in FIG. 6 may include an 40 encoding apparatus 300 shown in FIG. 3, and therefore, repeated description is omitted as appropriate. In addition, the transmitter 600 may further include a transmit unit 610, which is configured to allocate bits to a high frequency encoding parameter and a low frequency encoding parameter and a low frequency encoding parameter that are generated by the encoding apparatus 300, so as to generate a bit stream, and transmit the bit stream.

FIG. 7 is a block diagram that schematically shows a receiver 700 according to an embodiment of the present invention. The receiver 700 in FIG. 7 may include a decoding apparatus 500 shown in FIG. 5, and therefore, repeated description is omitted as appropriate. In addition, the receiver 700 may further include a receive unit 710, which is configured to receive an encoding signal for processing by the decoding apparatus 500.

In another embodiment of the present invention, a communications system is further provided, which may include a transmitter 600 that is described with reference to FIG. 6 or a receiver 700 that is described with reference to FIG. 7.

FIG. 8 is a schematic block diagram of an apparatus 60 according to another embodiment of the present invention. An apparatus 800 of FIG. 8 may be used to implement steps and methods in the foregoing method embodiments. The apparatus 800 may be applied to a base station or a terminal in various communications systems. In the embodiment of 65 FIG. 8, the apparatus 800 includes a transmitting circuit 802, a receiving circuit 803, an encoding processor 804, a decod-

**16** 

ing processor 805, a processing unit 806, a memory 807, and an antenna **801**. The processing unit **806** controls an operation of the apparatus 800, and the processing unit 806 may further be referred to as a Central Processing Unit (CPU). The memory 807 may include a read-only memory and a random access memory, and provides an instruction and data for the processing unit 806. A part of the memory 807 may further include a nonvolatile random access memory (NVRAM). In a specific application, the apparatus 800 may be built in a wireless communications device or the apparatus 800 itself may be a wireless communications device, such as a mobile phone, and the apparatus 800 may further include a carrier that accommodates the transmitting circuit 802 and the receiving circuit 803, so as to allow data transmitting and receiving between the apparatus 800 and a remote location. The transmitting circuit 802 and the receiving circuit 803 may be coupled to the antenna 801. Components of the apparatus 800 are coupled together using a bus system 809, where in addition to a data bus, the bus system 809 further includes a power bus, a control bus, and a status signal bus. However, for clarity of description, various buses are marked as the bus system 809 in a figure. The apparatus 800 may further include the processing unit 25 **806** for processing a signal, and in addition, further includes the encoding processor **804** and the decoding processor **805**.

The encoding method disclosed in the foregoing embodiments of the present invention may be applied to the encoding processor 804 or be implemented by the encoding processor 804, and the decoding method disclosed in the foregoing embodiments of the present invention may be applied to the decoding processor 805 or be implemented by the decoding processor 805. The encoding processor 804 or the decoding processor 805 may be an integrated circuit chip and has a signal processing capability. In an implementation process, steps in the foregoing methods may be completed by means of an integrated logic circuit of hardware in the encoding processor 804 or the decoding processor 805 or an instruction in a form of software. The instruction may be implemented or controlled by means of cooperation by the processor 806, and is used to execute the method disclosed in the embodiments of the present invention. The foregoing decoding processor may be a general purpose processor, a digital signal processor (DSP), an application-specific integrated circuit (ASIC), a field programmable gate array (FPGA) or another programmable logic component, a discrete gate or a transistor logic component, or a discrete hardware assembly, and can implement or execute methods, steps, and logical block diagrams disclosed in the embodiments of the present invention. The general purpose processor may be a microprocessor, and the processor may also be any conventional processor, decoder, and the like. Steps of the methods disclosed with reference to the embodiments of 55 the present invention may be directly executed and completed using a hardware decoding processor, or may be executed and completed using a combination of hardware and software modules in the decoding processor. A software module may be located in a mature storage medium in the art, such as a random access memory, a flash memory, a read-only memory, a programmable read-only memory, an electrically-erasable programmable memory, or a register. The storage medium is located in the memory 807, and the encoding processor 804 or the decoding processor 805 reads information from the memory 807, and completes the steps of the foregoing methods in combination with the hardware. For example, the memory 807 may store the obtained low

frequency encoding parameter for use by the encoding processor 804 or the decoding processor 805 during encoding or decoding.

For example, an encoding apparatus 300 in FIG. 3 may be implemented by the encoding processor **804**, and a decoding apparatus 500 in FIG. 5 may be implemented by the decoding processor 805.

In addition, for example, a transmitter 600 in FIG. 6 may be implemented by the encoding processor 804, the transmitting circuit **802**, the antenna **801**, and the like. A receiver <sup>10</sup> 700 in FIG. 7 may be implemented by the antenna 801, the receiving circuit 803, the decoding processor 805, and the like. However, the foregoing example is merely exemplary, and is not intended to limit the embodiments of the present 15 invention on this specific implementation manner.

Specifically, the memory 807 stores an instruction that enables the processor 806 and/or the encoding processor 804 to implement the following operations: dividing a to-beencoded time-domain signal into a low band signal and a 20 high band signal; performing encoding on the low band signal to obtain a low frequency encoding parameter; performing encoding on the high band signal to obtain a high frequency encoding parameter, and obtaining a synthesized high band signal according to the low frequency encoding 25 parameter and the high frequency encoding parameter; performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of a spectral envelope of the synthesized high band signal, a shape of a spectral envelope 30 of the short-time filtering signal is closer to a shape of a spectral envelope of the high band signal; and calculating a high frequency gain based on the high band signal and the short-time filtering signal. The memory 807 stores an instruction that enables the processor 806 or the decoding 35 processor 805 to implement the following operations: differentiating a low frequency encoding parameter, a high frequency encoding parameter, and a high frequency gain from encoded information; performing decoding on the low frequency encoding parameter to obtain a low band signal; 40 obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtering signal, where, compared with a shape of 45 a spectral envelope of the synthesized high band signal, a shape of a spectral envelope of the short-time filtering signal is closer to a shape of a spectral envelope of a high band signal; adjusting the short-time filtering signal using the high frequency gain to obtain a high band signal; and combining 50 the low band signal and the high band signal to obtain a final decoding signal.

The communications system or communications apparatus according to the embodiments of the present invention may include a part of or all of the foregoing encoding 55 apparatus 300, transmitter 600, decoding apparatus 500, receiver 700, and the like.

A person of ordinary skill in the art may be aware that, in combination with the examples described in the embodisteps may be implemented by electronic hardware or a combination of computer software and electronic hardware. Whether the functions are performed by hardware or software depends on particular applications and design constraint conditions of the technical solutions. A person skilled 65 in the art may use different methods to implement the described functions for each particular application, but it

**18** 

should not be considered that the implementation goes beyond the scope of the present invention.

It may be clearly understood by a person skilled in the art that, for the purpose of convenient and brief description, for a detailed working process of the foregoing system, apparatus, and unit, reference may be made to a corresponding process in the foregoing method embodiments, and details are not described herein again.

In the several embodiments provided in the present application, it should be understood that the disclosed system, apparatus, and method may be implemented in other manners. For example, the described apparatus embodiment is merely exemplary. For example, the unit division is merely logical function division and may be other division in actual implementation. For example, a plurality of units or components may be combined or integrated into another system, or some features may be ignored or not performed.

The units described as separate parts may or may not be physically separate, and parts displayed as units may or may not be physical units, may be located in one position, or may be distributed on a plurality of network units. Some or all of the units may be selected according to actual needs to achieve the objectives of the solutions of the embodiments.

The foregoing descriptions are merely specific implementation manners of the present invention, but are not intended to limit the protection scope of the present invention. Any variation or replacement readily figured out by a person skilled in the art within the technical scope disclosed in the present invention shall fall within the protection scope of the present invention. Therefore, the protection scope of the present invention shall be subject to the protection scope of the claims.

What is claimed is:

- 1. An encoding method for encoding a speech signal, comprising:
  - dividing the speech signal into a low band signal and a high band signal;
  - performing encoding on the low band signal to obtain a low frequency encoding parameter,
  - performing encoding on the high band signal to obtain a high frequency encoding parameter,
  - obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter;
  - performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtered signal; and
  - calculating a high frequency gain based on the high band signal and the short-time filtered signal.
- 2. The encoding method according to claim 1, wherein performing the short-time post-filtering processing on the synthesized high band signal comprises:
  - setting a coefficient of a pole-zero post-filter based on the high frequency encoding parameter, and
  - performing filtering processing on the synthesized high band signal using the pole-zero post-filter.
- 3. The encoding method according to claim 2, wherein ments disclosed in this specification, units and algorithm 60 performing the short-time post-filtering processing on the synthesized high band signal further comprises performing, using a first-order filter whose z-domain transfer function is  $H_{r}(z)=1-\mu z^{-1}$ , filtering processing on the synthesized high band signal that has been processed by the pole-zero postfilter after performing filtering processing on the synthesized high band signal using the pole-zero post-filter, and wherein μ is a preset constant or a value obtained by adaptive

calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

4. The encoding method according to claim 2, wherein performing encoding on the high band signal to obtain a high 5 frequency encoding parameter comprises:

performing, using a linear predictive coding (LPC) technology, encoding on the high band signal to obtain an LPC coefficient; and

using the LPC coefficient as the high frequency encoding parameter,

wherein a z-domain transfer function of the pole-zero post-filter is calculated using the following formula:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}},$$

and

wherein  $a_1, a_2, \dots a_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

- 5. The encoding method according to claim 1, further  $_{25}$ comprising generating an encoding bitstream according to the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain.
- **6**. A decoding method for decoding a speech signal, comprising:
  - differentiating a low frequency encoding parameter, a high frequency encoding parameter, and a high frequency gain from encoded information;
  - performing decoding on the low frequency encoding parameter to obtain a low band signal of the speech 35 signal;
  - obtaining a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter;
  - performing short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtered signal, wherein the shape of a spectral envelope of the short-time filtered signal is closer to a shape of a spectral envelope of the high band signal compared with the shape of a spectral envelope of the synthesized high band signal;

adjusting the short-time filtered signal using the high frequency gain to obtain a high band signal; and

combining the low band signal of the speech signal and 50 the high band signal to obtain a decoded signal.

7. The decoding method according to claim 6, wherein performing the short-time post-filtering processing on the synthesized high band signal comprises:

high frequency encoding parameter; and

performing filtering processing on the synthesized high band signal using the pole-zero post-filter.

8. The decoding method according to claim 7, wherein performing the short-time post-filtering processing on the 60 synthesized high band signal further comprises performing, using a first-order filter whose z-domain transfer function is  $H_{r}(z)=1-\mu z^{-1}$ , filtering processing on the synthesized high band signal that has been processed by the pole-zero postfilter after performing filtering processing on the synthesized 65 high band signal using the pole-zero post-filter, and wherein μ is a preset constant or a value obtained by adaptive

**20** 

calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

- **9**. The decoding method according to claim **7**, wherein the high frequency encoding parameter comprises:
  - a linear predictive coding (LPC) coefficient that is obtained by performing encoding using an LPC technology; and
  - a z-domain transfer function of the pole-zero post-filter is calculated using the following formula:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}},$$

and

15

wherein  $a_1, a_2 \dots a_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

- 10. An encoding apparatus for encoding a speech signal, comprising:
  - a memory that includes instructions:
  - at least one processor coupled to the memory and configured to receive the instructions, wherein when executing the instructions, the processor is configured to:

divide the speech signal into a low band signal and a high band signal;

perform encoding on the low band signal to obtain a low frequency encoding parameter;

perform encoding on the high band signal to obtain a high frequency encoding parameter; and

obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter, and

a filter coupled to the processor and configured to perform short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtered signal,

wherein when executing the instructions, the processor is further configured to calculate a high frequency gain based on the high band signal and the short-time filtered signal.

- 11. The encoding apparatus according to claim 10, 45 wherein the filter comprises a pole-zero post-filter configured to perform filtering processing on the synthesized high band signal, and wherein a coefficient of the pole-zero post-filter is set based on the high frequency encoding parameter.
- 12. The encoding apparatus according to claim 11, wherein the filter further comprises a first-order filter that is located behind the pole-zero post-filter and whose z-domain transfer function is  $H_r(z)=1-\mu z^{-1}$  and that is configured to perform filtering processing on the synthesized high band setting a coefficient of a pole-zero post-filter based on the 55 signal that has been processed by the pole-zero post-filter, and wherein µ is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.
  - 13. The encoding apparatus according to claim 11, wherein when executing the instructions, the processor is further configured to perform encoding on the high band signal using a linear predictive coding (LPC) technology to obtain an LPC coefficient, wherein the processor uses the LPC coefficient as the high frequency encoding parameter, wherein a z-domain transfer function of the pole-zero postfilter is calculated using the following formula:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}},$$

and wherein  $a_1, a_2 \dots a_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

14. The encoding apparatus according to claim 10, wherein when executing the instructions, the processor is 10 further configured to generate an encoding bitstream according to the low frequency encoding parameter, the high frequency encoding parameter, and the high frequency gain.

15. A decoding apparatus for decoding a speech signal, comprising:

a memory that includes instructions;

at least one processor coupled to the memory and configured to receive the instructions, wherein when executing the instructions, the at least one processor is configured to:

differentiate a low frequency encoding parameter, a high frequency encoding parameter, and a high frequency gain from encoded information;

perform decoding on the low frequency encoding parameter to obtain a low band signal of the speech 25 signal; and

obtain a synthesized high band signal according to the low frequency encoding parameter and the high frequency encoding parameter; and

a filter coupled to the processor and configured to perform short-time post-filtering processing on the synthesized high band signal to obtain a short-time filtered signal, wherein the shape of a spectral envelope of the short-time filtered signal is closer to the shape of a spectral envelope of a high band signal when compared with the shape of a spectral envelope of the synthesized high band signal, and

wherein when executing the instructions, the at least one processor is further configured to:

adjust the short-time filtered signal using the high frequency gain to obtain a high band signal; and combine the low band signal of the speech signal and the high band signal to obtain a decoded signal.

16. The decoding apparatus according to claim 15, wherein the filter comprises a pole-zero post-filter configured to perform filtering processing on the synthesized high band signal, and wherein a coefficient of the pole-zero post-filter is set based on the high frequency encoding parameter.

17. The decoding apparatus according to claim 16, wherein the filter further comprises a first-order filter that is located behind the pole-zero post-filter and whose z-domain transfer function is H<sub>z</sub>(z)=1-z<sup>-1</sup> and that is configured to perform filtering processing on the synthesized high band signal that has been processed by the pole-zero post-filter, and wherein μ is a preset constant or a value obtained by adaptive calculation that is performed according to the high frequency encoding parameter and the synthesized high band signal.

18. The decoding apparatus according to claim 16, wherein the high frequency encoding parameter is an LPC coefficient that is obtained using a linear predictive coding (LPC) technology, wherein a z-domain transfer function of the pole-zero post-filter is calculated using the following formula:

$$H_s(z) = \frac{1 - a_1 \beta z^{-1} - a_2 \beta^2 z^{-2} - \dots - a_M \beta^M z^{-M}}{1 - a_1 \gamma z^{-1} - a_2 \gamma^2 z^{-2} - \dots - a_M \gamma^M z^{-M}},$$

and wherein  $a_1, a_2, \dots a_M$  is the LPC coefficient, M is an order of the LPC coefficient, and  $\beta$  and  $\gamma$  are preset constants and satisfy  $0 < \beta < \gamma < 1$ .

\* \* \* \* \*